

COMPUTER TELEPHONY SERVER WITH IMPROVED FLEXIBILITY

TECHNICAL FIELD

This invention relates to telephony, and more particularly, to an improved computer telephony server that is capable of easily and conveniently interfacing with a variety of computer telephony environments, on a dynamic basis if necessary.

BACKGROUND OF THE INVENTION

The industry of computer telephony has experienced extensive growth over the last several years. One aspect of computer telephony involves software computer telephony applications which communicate over a local area network (LAN) or computer bus with a computer telephony (CT) server. The CT server then communicates with a private automated branch exchange ("PABX" or "PBX") for the purpose of implementing certain computer telephony functions. For example, "screen pop" is an example of a computer telephony application which may run on a personal computer (PC) and which may be implemented using a computer telephony server. In such an arrangement, a caller who dials in to a call center would have his telephone number read by a PBX. The telephone number is then forwarded through a local area network to a CT application program, which utilizes a table look-up in order to map the telephone number to a particular account number and account information. The account information is then read from a database and placed upon a computer screen to make it available to an agent for handling the incoming telephone call.

U.S. Patent No. 5,414,762 assigned to Q-sys describes a computer telephony server that can be implemented as a layer of software and assists with the interface between various PBXs and applications. Specifically, as described in the Q-sys patent, each of the variety of PBXs available on the market today may have differences in its command set, message structure and other features.

Accordingly, it is highly desirable that applications developers who write CT applications be able to write such applications independent of the particular PBX being utilized. Otherwise, each time an application is to work with a different PBX, the application would have to be rewritten.

5 The foregoing Q-sys patent solves the problem by offering a CT server that is capable of communicating with a variety of PBXs available on the market. The Q-sys telephony server has a standard message structure for communicating with applications over a local area network or a computer bus. Additionally, the arrangement taught by the Q-sys patent provides a means for translating the messages received from CT applications into the particular language and message structure of the particular PBX being utilized by the system.

10 The arrangement of the Q-sys patent provides a solution to the problem that applications developers used to face, namely, that each time their application was connected to a different PBX, it would have to be rewritten in order to understand and interpret the particular message structure, protocol, etc., of the PBX. By utilizing the Q-sys technique, a common language can be used by all applications, and a layer of independence is achieved between the CT application and the PBX.
15 Systems like that taught by the Q-sys patent were available nearly ten (10) years ago from Digital Equipment Corporation.

20 One problem with the Q-sys arrangement is that it does not take into account the fact that a PBX is only one particular type of telecommunications system capable of establishing a call between users. Specifically, the Q-sys system, while being capable of communicating with a variety of PBXs, is limited to situations where a PBX is the means by which arriving telephone calls are connected to users. In view of the foregoing, there exists a need in the art for an improved CT server which can operate in an environment where there are not only PBXs being utilized to establish calls between

users, but rather, where a variety of telephony systems (packet telephony networks, public switched telephony networks, etc.) may be utilized alternatively or simultaneously to construct such calls.

SUMMARY OF THE INVENTION

5 The foregoing and other problems of the prior art are overcome in accordance with the present invention, which relates to a method and apparatus for interfacing from computer telephony applications to a variety of computer telephony environments. In accordance with the present invention, a computer telephony server is utilized which provides a standard language for communicating with the call control mechanisms of a plurality of diverse telecommunications environments (such as PBXs, automatic call distributor (ACD) systems, Internet telephony environments, public switched telephone networks, etc.), and translating service requests and call status information between the standard instruction set utilized by the CT applications and the specific instruction set required by each telephony environment.

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 The server may be configured to communicate by means of the language and message structure of any of such diverse telecommunications environments through configuration instructions issued by the CT application operating on a remote computer. Alternatively, manual configuration of the server via human intervention may be used.

BRIEF DESCRIPTION OF THE DRAWINGS

20 Fig. 1 shows an architectural diagram of the exemplary embodiment of the present invention as used in a multiple telephony environment system; and

 Fig. 2 shows an exemplary embodiment of a data network telephony environment with which

computer telephony server 102 of Fig. 1 may work.

DETAILED DESCRIPTION OF THE PREFERRED EMBODIMENT

Fig. 1 shows an exemplary system in which the computer telephony server of the present invention may be used. The arrangement of Fig. 1 includes one or more applications 101, the inventive computer telephony server 102, and telephony environments 103 through 106. In the exemplary telephony environments shown, an Internet telephony environment 104 is included, as well as a PBX 105, the public switched telephone network (PSTN) 106, and an open switching system (103) built upon a CT media platform. The CT media platform is an open, general-purpose hardware and software platform suitable for the construction of media processing and switching systems. A representative CT media platform is available from the assignee of the present invention, and commercial implementations of switching systems built upon that platform are likewise available from their respective developers.

It is understood that the applications 101 may be any of a variety of computer telephony applications, such as screen pop applications, voice processing systems, etc. Other possible applications include call routing (determining to whom or to what system an incoming call should be routed) and database driven dialing, whereby an automated dialing application calls a predetermined set of telephone numbers listed in a database. Any one or more of these applications may require communications with CT server 102, as is conventional in the art.

In operation, information is exchanged between applications 101 and CT server 102, and between CT server 102 and one or more of the plurality of computer telephony environments 103 through 106. For purposes of explanation herein, and not limitation, the example of a screen pop

application 101 is utilized. In such an application, a telephone call is received through one of the telephony environments and delivered to the telephone instrument associated with a user of the screen pop application. Simultaneously, information about the arriving call is sent by the telephony environment (in its specific instruction set) to the CT server, which in turn forwards the information to the CT application (in the application's selected standard instruction set). The application discovers the caller's identity according to the existing computer telephony art by inspecting the automatic number identification (ANI) information or other user-entered data such as an account number. The CT application 101 uses this information to retrieve appropriate records from a business application database and presents the information to the user to expedite further interactions with the caller.

The information exchanged between the telephony server and the application relates to instructions which the application requires the computer telephony server to execute, as well as to status messages which are returned from the computer telephony environment (e.g., PBX 105) to the application 101. The computer telephony server 102 is arranged to be configurable for communicating with any of the variety of computer telephony environments. Additionally, such configuration may be done when the system is initially installed, or may be done on a dynamic basis, perhaps through commands issued by a particular application.

For example, an application may be capable of establishing telephony calls between its associated user and a remote party via a PBX or via the public switched telephone network. The application specifies, as part of its initial communication with CT server 102, the particular telephony environment desired. Computer telephony server 102 then translates all appropriate messages between application 101 and the selected telephony environment, 103 through 106. Such messages

are translated into appropriate commands and message structures which are applicable to either the PBX or the public network, as specified. The particular application may operate exactly the same regardless of whether the computer telephony environment is a PBX or a public network, or any of the other available environments. Additionally, if the application does not provide for such a selection, such a selection may be done automatically or in accordance with an operator program selection criteria and computer telephony server 102. Such an arrangement makes the application completely unaware of the commands and message structures of the particular telephony environment, and thus applications need not be rewritten each time the particular telephony environment is changed.

Fig. 2 shows a slightly more detailed diagram of the interconnection between the CT server 102 and a packet telephony network 104. As shown in Fig. 2, CT server 102 communicates with a gatekeeper 201 installed within the packet telephony network and configured according to the packet telephony art to supervise calls between the endpoints of interest to the CT application. The gatekeeper implements, for example, the H.323 protocols defined by the International Telecommunications Unit (ITU) for this purpose. Within the packet telephony environment, the gatekeeper 201 is responsible for supervising the establishment and disconnection of calls between endpoints within the packet telephony environment, such endpoints comprising both gateways 202 and 203 (points of entrance and egress from the packet telephony environment to other telephony environments such as the public switched telephone network) as well as telephone instruments 207 and 208 (connected directly to the packet telephony environment).

When so directed by one of the applications 101 (using the standard instruction set selected by the application), CT server 102 communicates over a data network with the gatekeeper 201 (using

the specific instruction set required by gatekeeper 201) requesting gatekeeper 201 to establish, disconnect, or take some other action with respect to calls within the gatekeeper's domain of supervision. Gatekeeper 201, in turn, utilizes the H.323 instructions specified by the ITU to implement the required operations within the packet telephony environment such as establishing internet telephony calls among end user telephones 205 through 208.

Note that the use of Internet telephony and gatekeeper 201 is entirely transparent to applications 101 and whether such telephony environment is used or whether the connection is set up through a PBX, CT media or public switched telephone network need not concern an application. Accordingly, the same application can be utilized together with a variety of telephony environments.

The above describes the preferred embodiment of the invention. It is understood that various modifications or additions will be apparent to those of skill in the art. Such additions are intended to be covered by the claims appended hereto.